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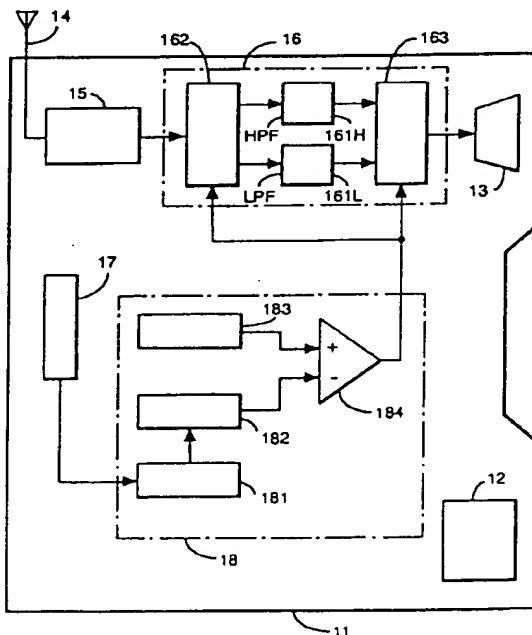
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(54) Telephone having a speech band limiting function.

(57) A telephone which includes a speech band switching circuit (16) selectively providing received speech with one of a plurality of frequency characteristics has a sensing circuit which senses the level of noise around the telephone, and a speech band control circuit (18) which compares the noise level sensed by the sensing circuit with a predetermined level and controls the speech band switching circuit on the basis of the result of comparison. When the noise level is higher than the predetermined level, the control circuit sends a control signal to the switching circuit. In response, the switching circuit selects a high-pass filter (161H) in place of a low-pass filter (161L). As a result, received speech is more easy to hear, despite noise surrounding the telephone, because the low frequency range is filtered out.

Fig.1.



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The present invention relates to a telephone capable of limiting the band-width of a received speech signal and, more particularly, though not exclusively, to a mobile telephone capable of making received speech more easy to hear, even in a noisy environment.

It has been customary to include, in a receiver circuit for a telephone, a filter for limiting the bandwidth of a received speech signal. The filter limits the received speech signal to a predetermined frequency band and thereby makes the speech more easy to hear. Specifically, the telephone has a casing and a transmitter (microphone) and a receiver (speaker) mounted on the casing. A radio section demodulates a signal received by an antenna. The filter mentioned above controls the frequency band of the output of the radio section. The radio section and the filter are connected to the receiver. In this configuration, the demodulated signal from the radio section has its frequency band limited by the filter, so that a speech signal is output from the receiver within the limited frequency band. Usually, the filter is of the kind which filters out a low frequency component in order to make received speech more easy to hear.

Various approaches have recently been proposed to improve the quality of speech received on a telephone. One of them is to provide a filter with a characteristic which enhances a low frequency component, by filtering out a high frequency component. Specifically, when a high frequency component is enhanced to promote more easy hearing, received speech sounds metallic and annoys the user of the telephone. In contrast, a filter of the kind filtering out a high frequency component allows received speech to sound soft, thereby improving the speech quality. Another approach is to allow the user to adjust the frequency of received speech on the telephone, to improve the ease of hearing, as taught in Japanese Patent Laid-Open Publication No. 1-123554 (document 1), by way of example.

However, a telephone with any of the conventional implementations for improving speech quality has the following problem. Assume that the telephone is used in a noisy urban environment outside of a building. Then, because the city noises, which are distributed generally, include a low frequency component, they are superposed on a speech signal whose frequency is limited to a low frequency component. The resulting speech output from the receiver is extremely hard to hear. For this reason, it has been customary, in consideration of operation in noisy environments, to limit received speech to a high frequency band, so as to promote more easy hearing, rather than to improve the quality of the speech.

The telephone proposed in the above mentioned document 1 promotes more easy hearing, because it allows the frequency characteristic of received speech to be adjusted in matching relation to the en-

vironment. However, the adjustment has to be made manually and, therefore, the user is forced to manipulate the telephone while conversing with the other party. Moreover, it is likely with such a manual control that noise drowns out speech for a moment, due to inadequate adjustment, and shuts off communication for that moment.

Features of a mobile telephone to be described below as an example are that it is capable of improving the quality of received speech and, in addition, of making it more easy to listen to the speech, even in a noisy urban environment.

In one arrangement, these features result from automatically changing the frequency band of the speech in matching relation to the noise level.

It is possible in a particular arrangement of a mobile telephone for an easy-to-hear condition to be attained adjustably at all times, without resorting to manual operation.

In a particular arrangement to be described below, a speech band switching circuit of a telephone selectively provides received speech with one of a plurality of frequency characteristics. A sensing circuit senses a noise level. A speech band control circuit compares the noise level sensed by the sensing circuit with a predetermined level, and controls the speech band switching circuit on the basis of the result of the comparison.

The speech band switching circuit may include a plurality of band-pass filters, each providing received speech with a particular frequency characteristic, and selectors for selecting one of the band-pass filters at a time. The sensing circuit may include a microphone mounted on the casing of the telephone for sensing the noise around the telephone.

A telephone having the above construction can automatically select the high frequency component or the low frequency component of received speech on the basis of the surrounding noise level. When the noise level is high, the telephone ensures that communication is possible by promoting ease of hearing, while sacrificing speech quality. For example, in a high noise level environment, the telephone selects the high-pass filter to filter out a low frequency component, thereby rendering the received speech more easy to hear. In a low noise level environment, the telephone selects the low-pass filter so as to filter out high frequency component and thereby causes the received speech to sound more soft, i.e. it improves the speech quality. Further, the telephone of an arrangement to be described determines whether or not the sound input to a transmitter is voice or noise, on the basis of the level of the sound, and selectively enables or disables the control circuit. The transmitter, therefore, plays the role of a noise detector at the same time.

The following description and drawings disclose, by means of examples, the invention which is charac-

terised in the appended claims, whose terms determine the extent of the protection conferred hereby.

In the drawings:-

Fig. 1 is a block schematic diagram showing an essential part of a mobile telephone incorporating one embodiment of the present invention; and
Fig. 2 is a block schematic diagram showing an essential part of an alternative embodiment of the present invention.

Referring to Fig. 1 of the drawings, a telephone, which is portable, is shown by way of example. The figure shows only a part of the telephone which is relevant to an understanding of the present invention. As shown, the telephone has a casing 11 on which a transmitter or microphone 12 and a receiver, or speaker, 13 are mounted. An antenna 14 is connected to a radio section 15 which is, in turn, connected to the receiver 13 via a speech band switching circuit 16. A signal received via the antenna 14 is demodulated by the radio section 15 and then applied to the speech band switching circuit 16. The switching circuit 16 switches the frequency band characteristic of the demodulated signal. As a result, the receiver 13 outputs received voice signals in the speech band selected by the switching circuit 16.

The speech band switching circuit 16 has a plurality of band-pass filters 161 connected in parallel, and selectors 162 and 163 respectively connected to the inputs and the outputs of the filters 161. The selectors 162 and 163 select one of the filters 161 at a time and connect it to the radio section 15 and the receiver 13. In the illustrative embodiment, the filters 161 are implemented as a high-pass filter (HPF) 161H and a low-pass filter (LPF) 161L for limiting the speech frequency to a high frequency band and a low frequency band, respectively. When the HPF 161H is selected, the high frequency components of the signal from the radio section 15 are passed through the HPF 161H and output as a speech signal via the receiver 13. The resulting voice output, having the high frequency component thereof enhanced, is more easy to hear. Conversely, when the LPF 161L is selected, the low frequency component of the speech signal is enhanced to improve the speech quality. The HPF 161H and LPF 161L may be implemented by filters customarily included in the tone control circuit of audio equipment.

The selectors 162 and 163 are switched by a control signal fed from a speech band control circuit 18 which will be described later. The selectors 162 and 163 may be implemented by mechanical or electrical switches. In the illustrative embodiment, the selectors 162 and 163 select the HPF 161H when a digital control signal goes low or select the LPF 161L when it goes high.

A microphone 17 is mounted on the rear of the casing 11 in order to sense the level of noise around the telephone. The microphone, or noise sense micro-

phone, 17 is connected to the speech band control circuit 18 which generates the above-mentioned control signal. The speech band control circuit 218 is made up of an adder 181, a mean unit 182, a store 183, and a comparator 184. The adder 181 adds consecutive noise signal levels sequentially sensed by the microphone 17. The mean unit 182 divides the output of the adder 181, i.e. the sum of the noise signal levels, by time to produce a mean noise signal level. The store 183 stores a predetermined reference noise signal level. The comparator 184 compares the mean noise signal level with the reference noise signal level stored in the store 183. When the mean level is higher than the reference level, the comparator 184 causes the digital control signal coupled to the selectors 162 and 163 to go low. When the former is lower than the latter, the comparator 184 causes the control signal to go high.

Specifically, the microphone 17 is made up of a microphone and a sampling circuit for sampling the output of the microphone, although not shown in the figure. The sampling circuit may be separately provided. The adder 181 sequentially adds the noise signal levels sampled by the sampling circuit. The mean unit 182 divides the sum of the noise signal levels by the number of times of sampling, so as to produce a mean noise signal level.

The reference noise signal level stored in the store 183 corresponds to a noise level in urban areas which would make the received speech hard to hear if the speech were passed through the LPF 161L. When the mean noise signal level is higher than the reference level, the speech band control circuit 18 determines that the noise around the telephone is too loud for the received speech to be heard if it is passed through the LPF 161L, thereby causing the control signal to go low. Conversely, when the former is lower than the latter, the circuit 18 determines that the noise is low enough for the received speech to be heard even if it is passed through the LPF 161L. As a result, the circuit 18 causes the control signal to go high.

When the control signal from the speech band control circuit 18 goes low, the selectors 162 and 163 select the HPF 161H. In this condition, the HPF 161H filters out the low frequency component of the signal coming in from the radio section 15 via the selector 162. The resulting voice from the receiver 13 is easy to hear despite the surrounding noise, although the quality is not improved. On the other hand, when the control signal goes high, the selectors 162 and 163 select the LPF 161L. The LPF 161L filters out the high frequency component of the incoming signal. Hence, the voice from the receiver 13 sounds soft, i.e., the quality is improved. Such a voice is more easily heard, because the surrounding noise level is low.

As stated above, the illustrative embodiment automatically selects the high frequency component or the low frequency component, depending on the level

of noise around the telephone. Specifically, when the noise is loud, the received speech signal is made easy to hear for better communication, although the speech quality is not improved. When the noise is not so loud, the speech quality is improved for more easy communication. This makes it needless for the user to adjust the frequency characteristic of received speech by hand. The telephone is, therefore, easy and convenient to operate and obviates the momentary interruption of communication due to any inadequate operation.

The microphone 17 is provided on the rear of the casing 11, i.e. on the opposite side to the transmitter 12. Hence, the voice of the user, speaking on the telephone, is substantially prevented from being input to the microphone 17 as noise. Preferably, the microphone 17 should be located at a position remote from the transmitter 12, e.g. at the rear of the receiver 13.

Assume that the transmitter and receiver are implemented as a hands-free adapter separate from a telephone casing. Then, the microphone 17 mounted on the casing will be sufficiently spaced apart from the transmitter and receiver and unlikely to be susceptible to the user's voice.

Generally, how easily a received voice can be heard depends on the person. In the light of this, a level adjuster may be connected between the store 183 and the comparator 184. Then, the user can adjust the reference noise signal level to be input to the comparator 184 beforehand, in matching relation to the auditory level particular to the user. In response, the telephone selects either the high frequency component or the low frequency component of the received speech signal by using the adjusted reference or threshold level, so that communication can be made under a condition matching the user's auditory level. Further, the speech band switching section 16 may be constructed such that the user can adjust the pass bands of the HPF 161H and LPF 161L to the user's taste.

Referring to Fig. 2, an alternative embodiment of the present invention will be described. Fig. 2, like Fig. 1, shows only part of the telephone relevant to the understanding of the present invention. Briefly, the alternative embodiment assigns the function of the microphone 17, Fig. 1, to a transmitter. As shown, the telephone has a casing 21, and a transmitter 22 and a receiver 23 mounted on the casing 21. An antenna is connected to a radio section 25 which is, in turn, connected to the receiver 23 via a speech band switching circuit 26. A signal received via the antenna 24 and demodulated by the radio section 25 is applied to the speech band switching circuit 26. In response, the switching circuit 26 selects either the high frequency component or the low frequency component of the speech signal, as in the previous embodiment.

Specifically, the speech band switching circuit 26 has a plurality of parallel band-pass filters 26a con-

nected to selectors 262 and 263 at opposite ends thereof. The selectors 262 and 263 select one of the filters 261 and connect it to the radio section 25 and the receiver 23. The filters 261 are implemented as an HPF 261H and an LPF 261L. When the selectors 262 and 263 select the HPF 261H, the HPF 261H enhances the high frequency component of the output of the radio section 25. As a result, a voice which is easy to hear is output from the receiver 23. When the selectors 262 and 263 select the LPF 261L, the low frequency component of the output of the radio section 25 is enhanced to improve the speech quality.

A speech band control circuit 28 delivers a digital control signal to the selectors 262 and 263. The selectors 262 and 263 select the HPF 261H when the control signal goes low, or select the LPF 261L when it goes high.

The speech band control circuit 28, similar to the circuit 18 of Fig. 1, and a voice detecting circuit 27 are connected to the transmitter 22. The voice detecting circuit 27 determines whether or not the user's voice has been input to the transmitter 22. The speech band control circuit 28 generates the above-mentioned control signal for operating the speech band switching circuit 26. A sampling circuit 29 samples noise signal levels sequentially detected at the transmitter 22. The speech band control circuit 28 has an adder 281, a mean unit 282, a store 283, and a comparator 284. The adder 281 adds the consecutive outputs of the sampling circuit 29. The mean unit 282 divides the sum of the noise signal levels by time to produce a mean noise signal level. The comparator 284 compares the mean noise signal level with a predetermined reference noise signal level stored in the store 283. The comparator 284 generates the control signal, which goes low if the mean signal level is higher than the reference signal level, or goes high if the former is lower than the latter.

The voice detecting circuit 27 has a sound discriminator 271 for determining the level of sound input to the transmitter 22, a store 272 storing a predetermined signal level, and a comparator 273 for comparing the detected sound level with the stored or reference signal level. When the sound level input to the transmitter is higher than the reference signal level, the circuit 27 determines that the user's voice has been input to the transmitter 22. If the former is lower than the latter, the circuit 27 determines that noise has been input to the transmitter 22. Because a voice has a particular spectrum, while noise has a flat spectrum, the voice detecting circuit 27 may be implemented as a circuit capable of discriminating voice and noise on the basis of a spectrum. Such an alternative circuit is well known in the speech recognition, synthesis and analysis art and will not be described specifically.

The output of the voice detecting circuit 27, indicating whether the input sound is voice or noise, is ap-

plied to a disabling circuit 285 in the speech band control circuit 28. The disabling circuit 285 may include a switch responsive to the output of the voice detecting circuit 27 for preventing the comparator 284 from supplying the control signal, and a holding circuit for holding the control signal when the supply of the control signal has been stopped. The output of the voice detecting circuit 27 thus disables the control circuit 28 when the input sound is voice, or enables it when the sound is noise. When the control circuit 28 is disabled by the output of the voice detecting circuit 27, the speech band switching circuit 26 continuously selects one of the HPF 261H and LPF 261L which it has selected immediately before. At this instant, which of the HPF 261H and LPF 261L is selected matters little, because no voice is usually output from the receiver 23 when a voice is input to the transmitter 22. When the control circuit 28 is enabled, meaning that noise around the telephone is input to the transmitter 22, the control circuit 28 controls the speech band switching circuit 26 in the same manner as in the previous embodiment. Specifically, the switching circuit 26 automatically selects the HPF 261H when the noise is loud or selects the LPF 261L if otherwise.

As stated above, this embodiment also automatically selects the high frequency component or the low frequency component, depending on the level of noise around the telephone. Specifically, when the noise is loud, the received speech signal is made easy to hear for better communication, although the speech quality is not improved. When the noise is not so loud, the speech quality is improved for more easy communication. This makes it needless for the user to adjust the frequency characteristic of the received voice by hand. The telephone is, therefore, easy and convenient to operate and obviates the momentary interruption of communication due to inadequate operation of the control. In addition, because the transmitter 22 plays the role of a noise microphone at the same time, it is not necessary to add a microphone to an existing telephone. The telephone may therefore be miniature and light weight. This embodiment, like the previous embodiment, is practicable even with a telephone having a hands-free adapter.

In summary, it will be seen that the arrangements described provide a telephone capable of sensing the level of noise around it, comparing it with a predetermined signal level, and switching the frequency characteristic of received speech on the basis of the noise level. Therefore, the telephone can automatically select the high frequency component or the low frequency component of received speech on the basis of the noise level. When the noise level is high, the telephone ensures better communication by promoting ease of hearing while sacrificing speech quality. When the noise level is low, the telephone ensures easy communication by improving speech quality. More specifically, in a high noise level environment,

the telephone selects a high-pass filter to filter out a low frequency component, thereby rendering received speech easy to hear. In a low noise level environment, the telephone selects a low-pass filter so as to filter out a high frequency component and thereby causes received speech to sound soft, i.e. it improves the speech quality. This makes it needless for the user to adjust the frequency characteristic of the received speech by hand. As a result, the telephone is easy and convenient to operate and is free from the momentary shut-off of reception due to any erroneous operation.

Further, the telephone which has been described above determines whether or not sound input to a transmitter is voice or noise on the basis of the level of the sound, and selectively enables or disables a speech band control circuit. The transmitter, therefore, plays the role of a noise microphone at the same time and makes it needless to add an extra microphone to the telephone. This also reduces the size and weight of the telephone.

It will be understood that, although particular embodiments have been described, by way of example, variations and modifications thereof, as well as other embodiments, may be made within the scope of the protection sought, as defined by the appended claims.

Claims

1. A telephone including a speech band switching circuit (16,26) for selectively providing received speech with one of a plurality of frequency characteristics, a sensor (17,27) for sensing a noise level, and a control circuit (18,28) for comparing the noise level sensed by the sensor (17,27) with a predetermined signal level, and for controlling the speech band switching circuit (16,26) on the basis of the result of the comparison.
2. A telephone including a plurality of band-pass filters (161L, 161H) each providing a receive speech signal with a particular frequency characteristic, selectors (162,163) for selecting one of the plurality of band-pass filters at a time, a microphone (17) mounted on a casing (11) of the telephone for sensing noise around the telephone, a store (183) storing a predetermined signal level, and a comparator (184) for comparing a noise level sensed by the microphone (17) with the signal level stored in the store (183), wherein the selectors (162,163) are operated by an output of the comparator (184).
3. A telephone including a plurality of band-pass filters (261L,261H) each for providing a received speech signal with a particular frequency charac-

teristic, selectors (262,263) for selecting one of the plurality of band-pass filters (261L,261H) at a time, a voice detecting circuit (27) for determining, based on a sound level input to a transmitter (22) of the telephone, whether or not a speech signal to be transmitted has been input to the transmitter (22), a store (283) storing a predetermined signal level, and a comparator for comparing the level of the sound input to the transmitter (22), and containing noise, with the signal level stored in the store (283), wherein the selectors (262,263) are operated according to an output of the voice detecting circuit (27) and an output of the comparator (284).

4. A telephone as claimed in claim 1, wherein the speech band switching circuit (26) includes a high-pass filter (161H), a low-pass filter (161L), and selectors (162,163) for selecting one of the high-pass filter (161H) and the low-pass filter (161L), and wherein the selectors (162,163) select the high-pass filter (161H) when the noise level is higher than the predetermined signal level, and select the low-pass filter (161L) when the noise level is lower than the predetermined signal level.
5. A telephone as claimed in claim 1, wherein the control circuit (18) includes an adder (181) for adding noise signal levels sequentially sensed by the sensor (17), a mean unit (182) for dividing the sum of the noise signal levels by time to produce a mean value, and a comparator (184) for comparing the mean value with the predetermined signal level.
6. A telephone as claimed in any one of the preceding claims which is a mobile telephone.

Fig.1.

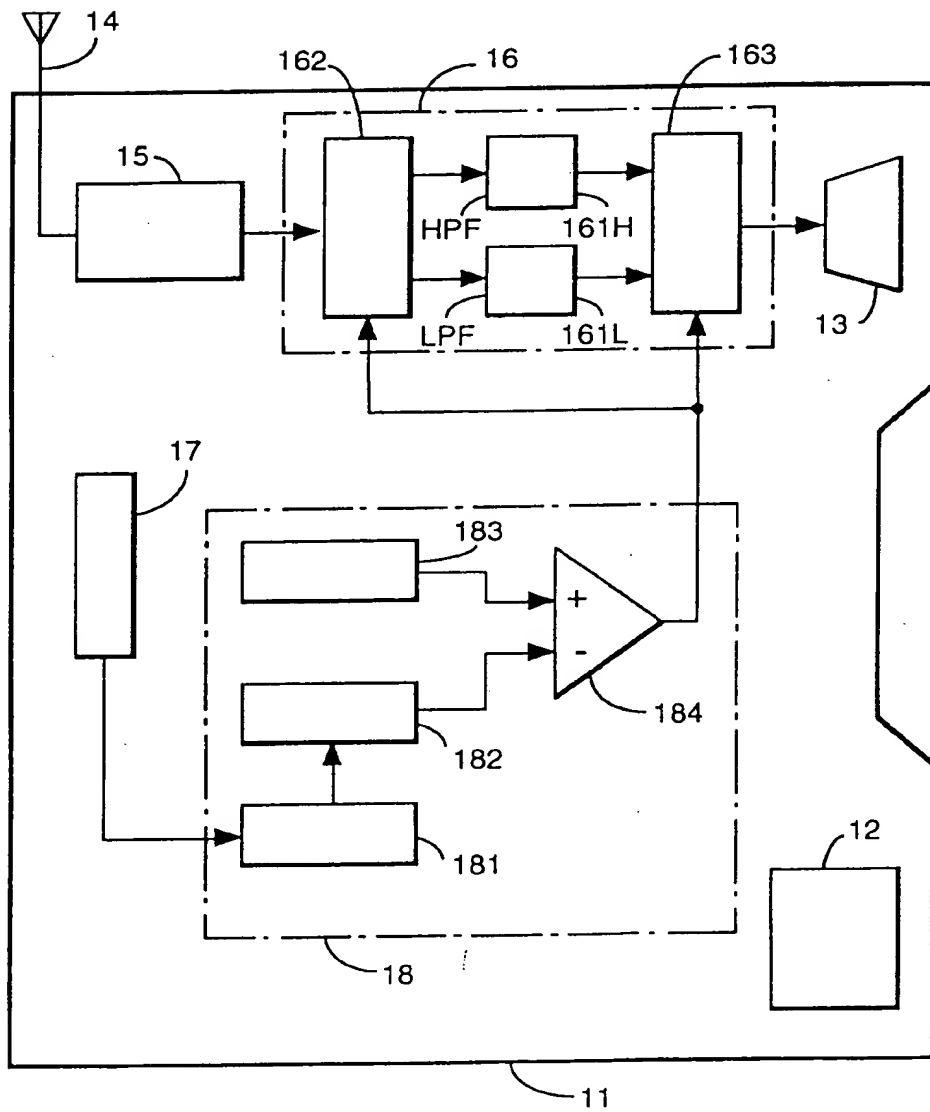
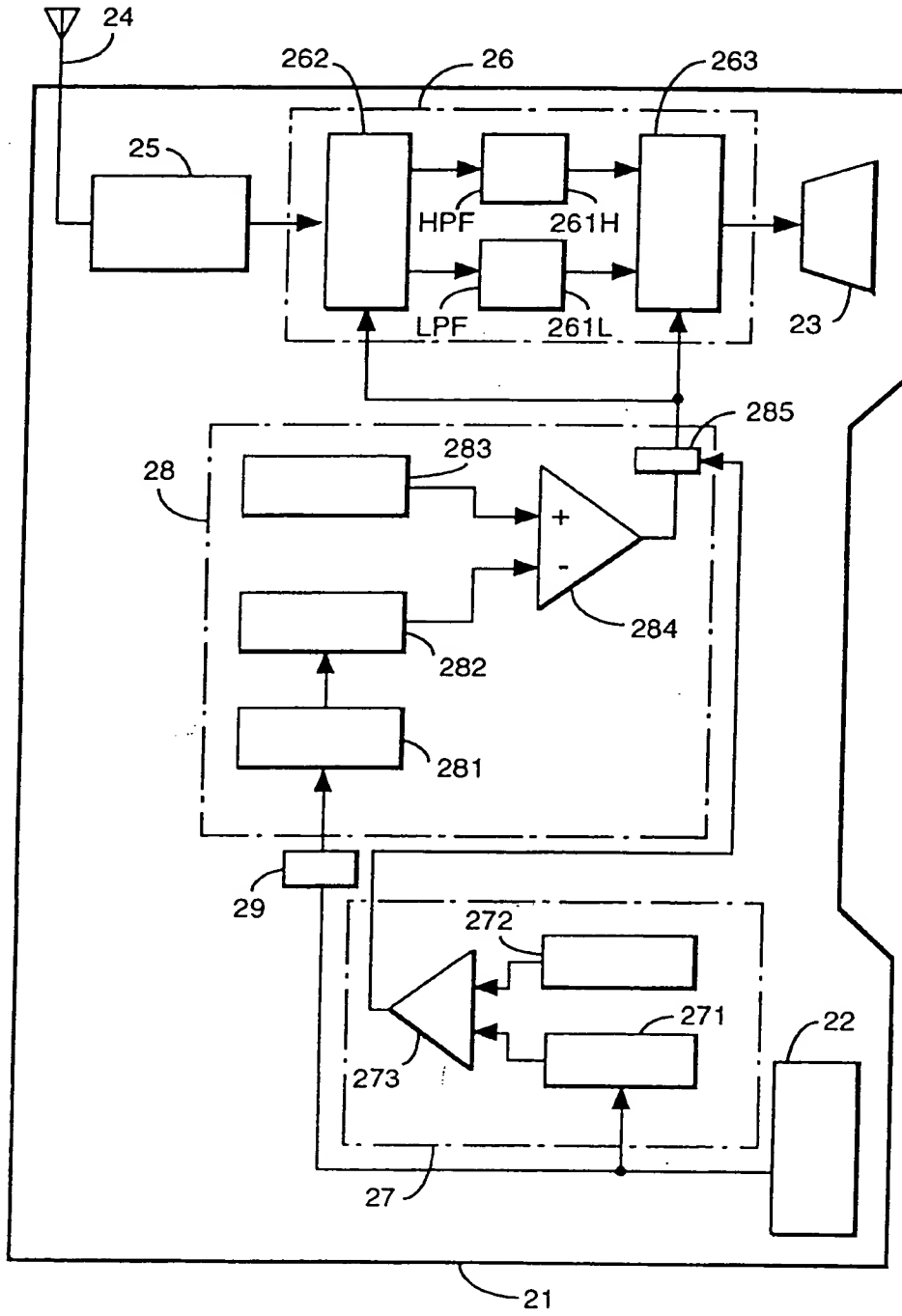


Fig.2.



European Patent
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EUROPEAN SEARCH REPORT

Application Number

DOCUMENTS CONSIDERED TO BE RELEVANT			EP 95302011.2
Category	Citation of document with indication, where appropriate, of relevant passages	Relevant to claim	CLASSIFICATION OF THE APPLICATION (Int. Cl. 6)
A	<u>WO - A - 93/06 679</u> (MITEL) * Abstract; page 1, line 1 - page 3, line 7; fig. 1; claim 1 * --	1, 2	H 04 M 1/00
A	<u>EP - A - 0 222 083</u> (IBM) * Abstract; column 1, line 4 - column 3, line 17; fig. 1; claims 1, 3 * --	1, 2	
A	<u>EP - A - 0 343 792</u> (NOKIA) * Abstract; column 1, lines 1-54; fig. 1; claims 1, 2 * --	1, 2, 6	
A	<u>EP - A - 0 477 158</u> (ERICSON) * Abstract; column 1, line 3 - column 3, line 54; fig. 1, 2; claims 1, 7, 17 * ----	1, 2, 6	
			TECHNICAL FIELDS SEARCHED (Int. Cl. 6) H 04 M G 10 L H 03 H H 04 B
The present search report has been drawn up for all claims			
Place of search VIENNA	Date of completion of the search 30-06-1995	Examiner BADICS	
CATEGORY OF CITED DOCUMENTS X : particularly relevant if taken alone Y : particularly relevant if combined with another document of the same category A : technological background O : non-written disclosure P : intermediate document T : theory or principle underlying the invention E : earlier patent document, but published on, or after the filing date D : document cited in the application L : document cited for other reasons & : member of the same patent family, corresponding document			

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